DOCUMENT CONTROL DATA - R & D

Security classification of title, body of abstract and indexing annotation must be entered when the overall report is classified

1. ORIGINATING ACTIVITY (Corporate author)	2a. REPORT SECURITY CLASSIFICATION
Cooley Electronics Laboratory University of Michigan Ann Arbor, Michigan 48105	Unclassified
	2b. GROUP

3. REPORT TITLE

MODULATION BY LINEAR-MAXIMAL SHIFT-REGISTER SEQUENCES: AMPLITUDE, BIPHASE AND COMPLEMENT-PHASE MODULATION

4. DESCRIPTIVE NOTES (Type of report and inclusive dates)

C.E.L. Technical Report No. 216 -- December 1971

5. AUTHOR(S) (First name, middle initial, last name)

Theodore G. Birdsall
Richard M. Heitmeyer
Kurt Metzger

6. REPORT DATE	7a. TOTAL NO. OF PAGES	7b. NO. OF REFS
December 1971	40	4

8a. CONTRACT OR GRANT NO.	9a. ORIGINATOR'S REPORT NUMBER(S)
N00014-67-A-0181-0032	036040-9-T
b. PROJECT NO.	
c.	9b. OTHER REPORT NO(S) (Any other numbers that may be assigned this report)
d.	TR216

10. DISTRIBUTION STATEMENT

Approved for public release; distribution unlimited.

11. SUPPLEMENTARY NOTES	12. SPONSORING MILITARY ACTIVITY
	Office of Naval Research Department of the Navy Arlington, Va. 22217

13. ABSTRACT

This report describes the use of linear-maximal shift-register sequences either to amplitude modulate (AM) or to angle modulate a sinusoidal carrier. Two cases of angle modulation are considered: biphase modulation (BM) and complement-phase modulation (CM). In biphase modulation the carrier angle switches between $+\pi/2$ and $-\pi/2$, whereas in complement-phase modulation the carrier angle switches between $+\pi/4$ and $-\pi/4$. It is shown that the ratio of the average power to the peak power for AM is only one-half as large as for either BM or CM. On the other hand, the ratio of the carrier power to the average power for either AM or CM is one-half, but for BM this ratio is approximately zero. Finally, a replica-correlation technique for use with any of the three modulations is described.

Best
Available
Copy

14. KEY WORDS	LINK A		LINK B		LINK C	
	ROLE	WT	ROLE	WT	ROLE	WT
Linear-maximal shift register sequences						
Amplitude modulation						
Biphase modulation						
Complement-phase modulation						
Replica-correlation technique						

Technical Report No. 216

036040-9-T

MODULATION BY LINEAR-MAXIMAL SHIFT REGISTER SEQUENCES:

AMPLITUDE, BIPHASE AND

COMPLEMENT-PHASE MODULATION

by

Theodore G. Birdsall
Richard M. Heitmeyer
Kurt Metzger

COOLEY ELECTRONICS LABORATORY

Department of Electrical and Computer Engineering
The University of Michigan
Ann Arbor, Michigan

for

December 1971

ABSTRACT

This report describes the use of linear-maximal shift-register sequences either to amplitude modulate (AM) or to angle modulate a sinusoidal carrier. Two cases of angle modulation are considered: biphase modulation (BM) and complement-phase modulation (CM). In biphase modulation the carrier angle switches between $+\pi/2$ and $-\pi/2$, whereas in complement-phase modulation the carrier angle switches between $+\pi/4$ and $-\pi/4$. It is shown that the ratio of the average power to the peak power for AM is only one-half as large as for either BM or CM. On the other hand, the ratio of the carrier power to the average power for either AM or CM is one-half, but for BM this ratio is approximately zero. Finally, a replica-correlation technique for use with any of the three modulations is described.

iii

ACKNOWLEDGMENT

TABLE OF CONTENTS

LIST OF ILLUSTRATIONS

LIST OF TABLES

1. INTRODUCTION

A common technique for sounding the multipath structure of an acoustical or an electromagnetic channel is to transmit a periodic signal consisting of a sequence of rectangular pulses, modulating a carrier. The implementation of this technique, however, often leads to the following problem.

If d is the duration of each of the rectangular pulses, then d must be kept small in order to resolve the different arrivals from each transmission path. On the other hand, if T denotes the period of the transmission, then T must be kept large so that all of the arrivals from one transmitted pulse can be distinguished from those of the succeeding pulse. Thus, the nature of the propagation mechanism places an upper bound on the duty cycle d/T. Furthermore, the equipment used to transmit the signal may also place an upper bound on the signal's peak power P so that the quantity $P(d/T)$ may be bounded from above. However, $P(d/T)$ is the average power in the signal, and if this quantity is not large enough to compensate for both the attenuation in the channel, and the channel and system noise, then this technique cannot be successfully implemented.

Over the years, however, various pulse-compression techniques suitable for periodic transmissions have been developed to circumvent this basic problem. The idea behind these techniques is

-1-

to construct a signal that has the same peak power, and effectively the same duty cycle, as the repeated pulse signal but with a larger average power. One such technique, which has gained in popularity since the advent of digital circuitry, involves the use of linear-maximal shift-register (LMSR) sequences to modulate the carrier. The transmitted signal is constructed to have a sufficiently large average-power to peak-power ratio while, at the receiver, the reception is correlated with a replica in order to achieve the effect of a small duty cycle.

This report describes three different LMSR modulation techniques that have been used in the MIMI (University of Michigan - University of Miami) underwater sound propagation experiments. Section 2 briefly reviews the linear-maximal, shift-register sequences. Section 3 describes the modulation techniques themselves and derives the power spectra for each case. In Section 4, a replica-correlation technique for achieving the effect of a small duty cycle is described. In large part the material contained in this report is taken from "MIMI Processing Techniques" (Ref. 1).

2. A REVIEW OF LINEAR-MAXIMAL, SHIFT-REGISTER SEQUENCES

A LMSR sequence $\{m_k\}$ is a periodic, binary sequence $m_k \in \{0, 1\}$, which can be generated by certain shift-register generators using mod-2 arithmetic. Methods of generating these sequences and their algebraic properties have been studied extensively (Refs. 2 and 3). Here only the following three properties are presented:

1. Periodic Property

The period L of a LMSR sequence is of the form

$$L = 2^n - 1 \qquad n = 1, 2, \ldots$$

The number n can be associated with the number of stages in the shift-register generator.

2. Balance Property

The total number of 1's in one period of an LMSR sequence is

$$(L+1)/2 = 2^{n-1}$$

and the total number of 0's is

$$(L-1)/2 = 2^{n-1} - 1$$

3. Pseudo-Random Property

The occurrence of 1's and 0's in an **LMSR** sequence has the appearance of having been generated by successive trials of a fair-coin tossing experiment. (A more precise statement of the pseudo-random property can be found in Ref. 2.)

The above three properties state (i) that **LMSR** sequences can only have periods of the form $L = 2^n - 1$, (ii) that there are approximately an equal number of 1's and -1's in an **LMSR** sequence, and (iii) that the order in which the 1's and 0's occur does not follow a simple pattern. A specific example of an **LMSR** sequence from a 4-stage generator, $n = 4$, with period, $L = 2^4 - 1 = 15$, is

$$\ldots 1\ 1\ 1\ 1\ 0\ 0\ 0\ 1\ 0\ 0\ 1\ 1\ 0\ 1\ 0 \ldots .$$

In order to describe the use of **LMSR** sequences in modulation schemes, it is convenient to introduce the notion of the associated binary waveform. Basically, the idea is to construct a periodic, binary-valued function of time, $m(t)$, by associating logical 1's in $\{m_k\}$ with $+1$ values of $m(t)$ and logical 0's in $\{m_k\}$ with -1 values of $m(t)$. Specifically, we define

$$m(t) = \begin{cases} +1 & \text{for} \quad t \in [kd, (k+1)\,d) \quad \text{if} \quad m_k = +1 \\ -1 & \text{for} \quad t \in [kd, (k+1)\,d) \quad \text{if} \quad m_k = -1 \end{cases}$$

The number d is known as the digit duration and the period of m(t) is easily seen to be equal to

$$T = Ld$$

The binary waveform associated with the sequence $\{m_k\}$ illus-trated above is shown in Fig. 1.
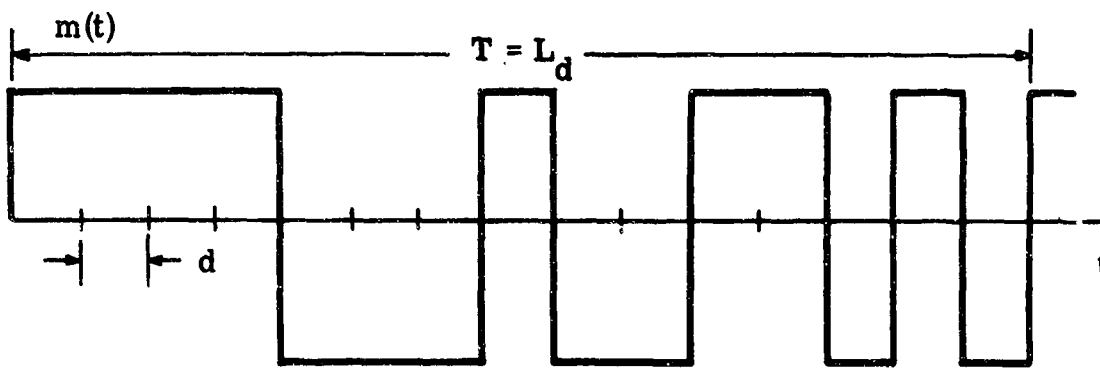

Fig. 1. One period of a LMSR binary waveform

The utility of LMSR waveforms as modulation waveforms is due to the form of the autocorrelation function

$$R_m(\tau) = \frac{1}{T} \int_0^T m(t)\, m^*(t+\tau)\, dt \qquad (1)$$

and the corresponding power spectra

$$S_m(f) = \int_{-\infty}^{\infty} R_m(\tau) e^{-j2\pi f} d\tau \qquad (2)$$

In Ref. 2 it is shown that $R_m(\cdot)$ is periodic with period T and for $\tau \in [-T/2, T/2]$,

$$R_m(\tau) = \begin{cases} 1 - \dfrac{(1+L)}{L} \quad \dfrac{|\tau|}{d} & |\tau| \leq d \\[4mm] -\dfrac{1}{L} & \text{otherwise} \end{cases}$$

Moreover, the power spectra for $m(t)$ is a line spectra given by,

$$S_m(f) = A(f) P(f) \left[\sum_n \delta(f - n/T) \right] \qquad (3)$$

where

$$P(f) = \left[\frac{\sin \pi df}{\pi df} \right]^2$$

and

$$A(f) = \begin{cases} 1/L^2 & f = 0 \\[4mm] (1+L)/L^2 & f \neq 0 \end{cases}$$

Illustrations of $R_m(\tau)$ and the root-mean-squared (RMS) power spectrum, $[S_m(f)]^{\frac{1}{2}}$, appear in Fig. 2.
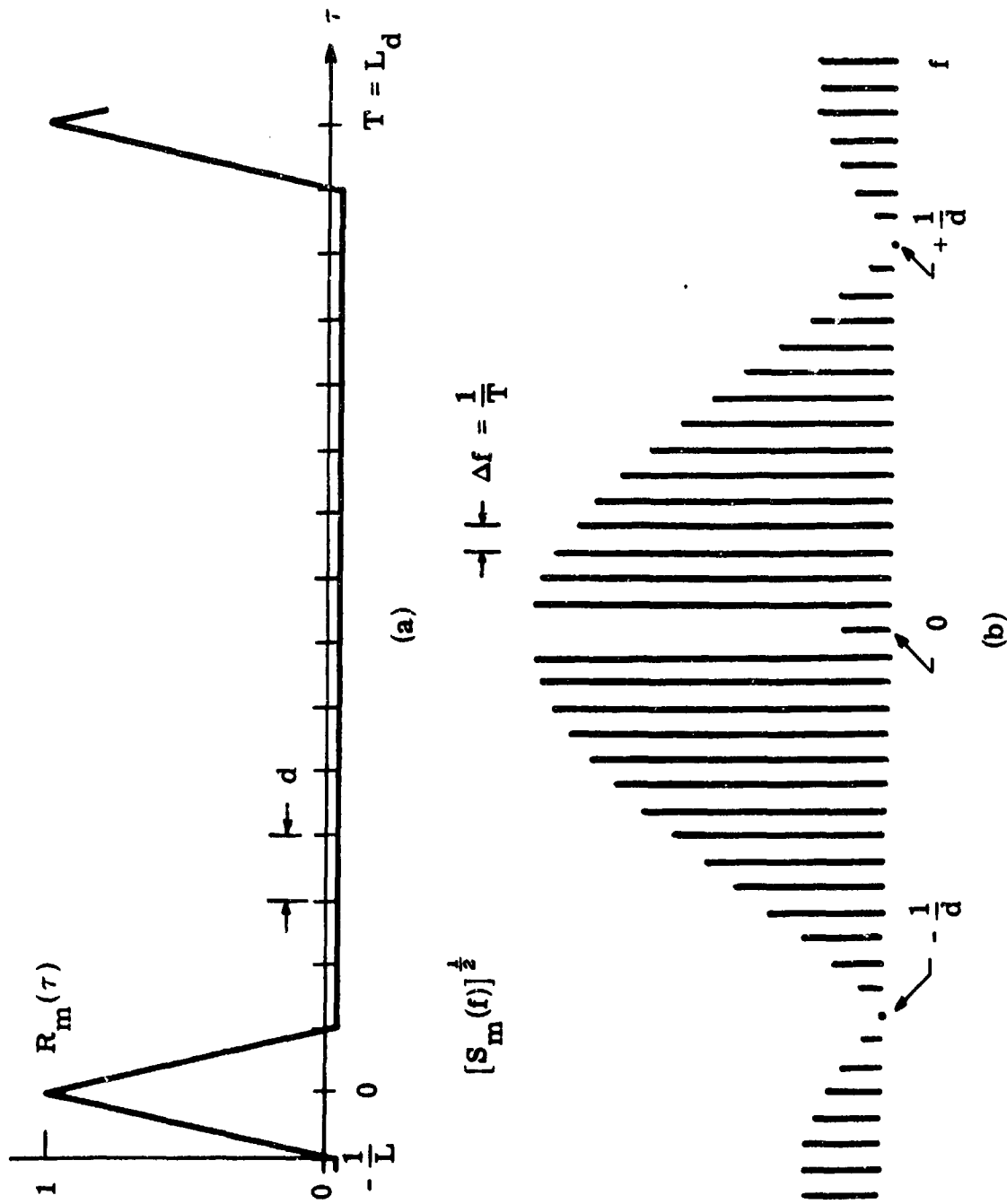
Fig. 2.   (a) Autocorrelation function $R_m(\tau)$,
(b) RMS power spectrum $[S_m(f)]^{\frac{1}{2}}$

3. AMPLITUDE MODULATION, BIPHASE MODULATION

AND COMPLEMENT-PHASE MODULATION

The basic idea behind LMSR modulation is to cause either the amplitude or the phase of a carrier to switch between two values depending on the value of the modulating waveform $m(t)$. In the case of amplitude modulation (AM), the amplitude switches between $+1$ and 0, in biphase modulation (BM), the phase switches between $+\pi/2$ and $-\pi/2$ and in complement-phase modulation (CM), the phase switches between $+\pi/4$ and $-\pi/4$. More precisely, if $\hat{s}(t)$ is the low-pass waveform associated with the modulated signal $s(t)$,

$$s(t) = \mathrm{Re}\left[\hat{s}(t)\left(\sqrt{2}\; e^{-j2\pi f_c t}\right)\right] \tag{4}$$

then for AM

$$s_a(t) = \{[1 + m(t)]/2\}\sqrt{2}\;\sin(2\pi f_c t)$$

$$\hat{s}_a(t) = [1 + m(t)]\;j/2 \tag{5}$$

and for BM

$$s_b(t) = m(t)\sqrt{2}\;\sin 2\pi f_c t = \sqrt{2}\;\cos\left[2\pi f_c t - m(t)\;\pi/2\right]$$

$$\hat{s}_b(t) = m(t)\;j \tag{6}$$

and for CM

-8-

$$s_c(t) = \sqrt{2} \cos \left[2\pi f_c t - m(t)\, \pi/4 \right]$$

(7)

$$\hat{s}_c(t) = \left[1 + m(t)\, j \right] / \sqrt{2}$$

In each of the above cases, it is assumed that there is an integral number of cycles of carrier for each digit in $m(t)$. If this number is denoted by D, then $s(t)$ is periodic with period

$$T = dL$$

where

$$d = D/f_c$$

The complex low-pass signals for each of the three cases are illustrated in Fig. 3 and portions of the modulated waveforms appear in Fig. 4.

In order to obtain the power spectra for the three different modulation schemes, the autocorrelation functions for the low-pass signals must first be determined. These functions may be determined from the definition of the autocorrelation of a complex signal (Eq. 1) by a straightforward computation. The results are,

for AM

$$R_a(\tau) = \left[(1 + 2/L) + R_m(\tau) \right] / 4$$

(8)

for BM

$$R_b(\tau) = R_m(\tau)$$

(9)

$\hat{s}(t)$ for $m(t) = +1$ $\hat{s}(t)$ for $m(t) = +1$ $\hat{s}(t)$ for $m(t) = 1$

$\hat{s}(t)$ for $m(t) = -1$

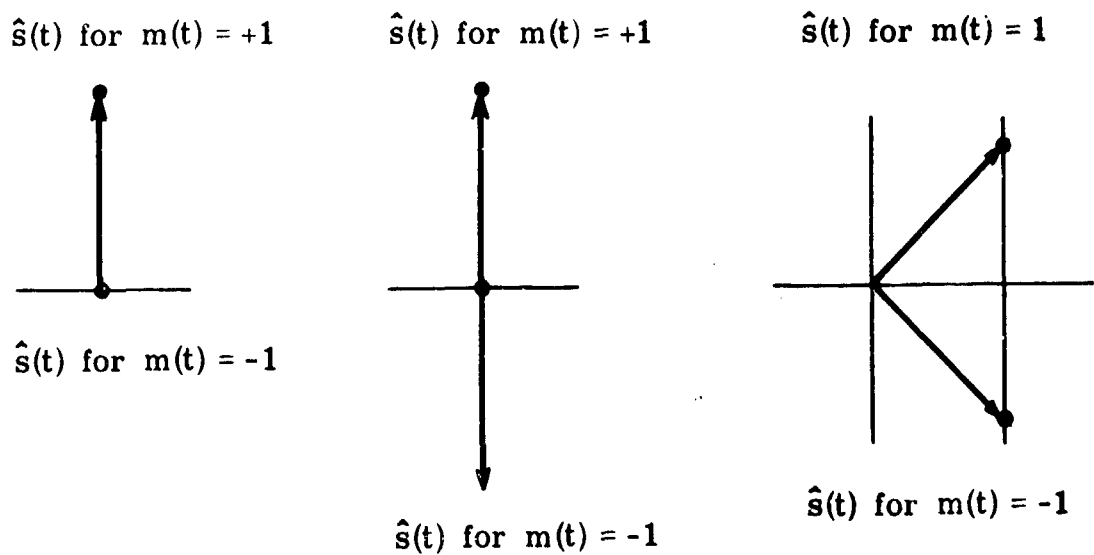$\hat{s}(t)$ for $m(t) = -1$ $\hat{s}(t)$ for $m(t) = -1$

Fig. 3. The low-pass waveform $\hat{s}(t)$;
(a) AM,   (b) BM,   (c) CM

and for CM

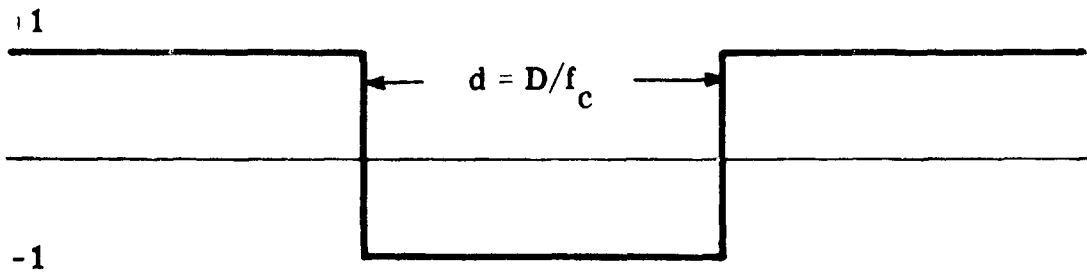$$R_c(\tau) = [1 + R_m(\tau)]/2 \tag{10}$$

Next, the spectra for the low-pass signals are obtained as the Fourier

transforms of the corresponding autocorrelation functions. We have,
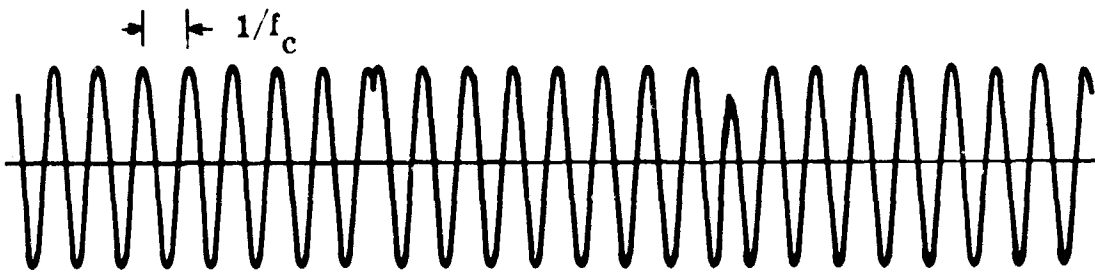
for AM

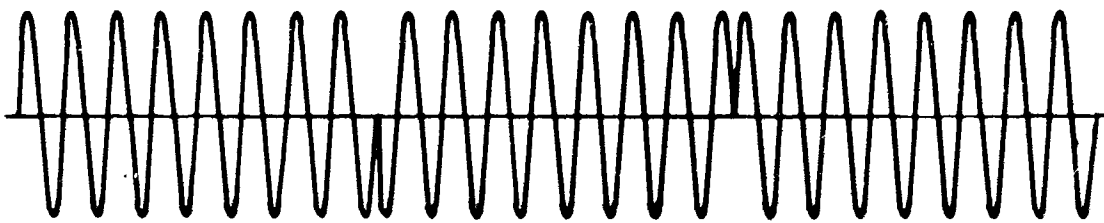$$S_a(f) = [(1 + 2/L)\,\delta(f) + S_m(f)]/4 \tag{11}$$

for BM

$$S_b(f) = S_m(f) \tag{12}$$

Fig. 4. The modulated signal s(t); (a) a portion of m(t) , (b) CM, (c) BM, (d) AM

and for CM

$$S_c(f) = [\delta(f) + S_m(f)]/2 \tag{13}$$

or alternatively, if $\hat{S}(f)$ denotes either $S_a(f)$, $S_b(f)$ or $S_c(f)$, then

$$\hat{S}(f) = A(f) P(f) \sum_n (f - n/T) \tag{14}$$

where again

$$P(f) = [\sin \pi df/\pi df]^2$$

and $A(f)$ is given in Table I for the three cases.

Table I. The function $A(f)$ for **AM, BM and CM**

AM	$A(f) = \dfrac{1}{4} \dfrac{(1 + L)^2}{L^2}$	$f = 0$
	$A(f) = \dfrac{1}{4} \dfrac{1 + L}{L^2}$	$f \neq 0$
BM	$A(f) = 1/L^2$	$f = 0$
	$A(f) = \dfrac{1 + L}{L^2}$	$f \neq 0$
CM	$A(f) = \dfrac{1 + L^2}{2L^2}$	$f = 0$
	$A(f) = \dfrac{1 + L}{2L^2}$	$f \neq 0$

To find the power spectrum for the modulated waveform, $S(f)$, we note that if the number of cycles per digit, D, is larger than one, then the first zero of $P(f)$, $d^{-1} = f_c/D$, is less than f_c. Thus, $s(t)$ is essentially bandlimited so that,

$$S(f) = \frac{1}{2} [\hat{S}(f - f_c) + \hat{S}(f + f_c)] \tag{15}$$

Furthermore, the total average power in the modulated signal, P_s, can be obtained from the autocorrelation function of the low-pass signal by noting that,

$$P_s = \int_{-\infty}^{\infty} S(f)\, df$$

$$= \frac{1}{2} \int_{-\infty}^{\infty} [\hat{s}(f - f_c) + S(f + f_c)]\, df$$

$$= \frac{2}{2} \int_{-\infty}^{\infty} \hat{S}(f)\, df$$

or

$$P_s = \hat{R}(0) \tag{16}$$

Moreover, it is easy to show that the power in the carrier frequency line, P_c, is given by,

$$P_c = A(0) \tag{17}$$

and that the peak RMS power, P, is

$$P = \max\left[\hat{s}(t)\right] = 1 \tag{18}$$

Finally, the above equations (Eqs. 16, 17 and 18) may be used to determine the ratio of the total average power to the peak power,

$$\alpha = P_s/P = \hat{R}(0) \tag{19}$$

and the ratio of the carrier power to the total average power,

$$\beta = P_c/P_s = A(0)/\hat{R}(0) \tag{20}$$

for the three different cases. The results are summarized in Table II.

Table II. The ratios α and β for AM, BM and CM

	$\alpha = P_s/P$	$\beta = P_c/P_s$
AM	$\frac{1}{2}\left(1 + \frac{1}{L}\right) \approx \frac{1}{2}$	$\frac{1}{2}\left(1 + \frac{1}{L}\right) \approx \frac{1}{2}$
BM	1	$\frac{1}{L^2} \approx 0$
CM	1	$\frac{1}{2}\left(1 + \frac{1}{L^2}\right) \approx \frac{1}{2}$

In the above paragraphs it has been seen that the power spectra for each of the three modulation schemes has the basic $[(\sin x)/x]^2$

form except at the carrier frequency line. The difference in the three modulation schemes appears in the ratios α and β. For a fixed peak power, the average power in the AM signal is only about half of the average power in either the BM or CM signal. On the other hand, both the AM and CM signal have approximately half of the total power contained in the carrier line, whereas the BPM signal has essentially no power contained in the carrier line.

The choice of which modulation scheme to use in a particular setting depends, of course, not only on the quantities α and β, but on the ease with which the modulation is implemented. Of the three types, AM modulation is the easiest to implement since it only involves gating the carrier on or off depending on the bits in the LMSR sequence. If more signal power is needed, however, the choice between BM and CM can be made on the basis of whether or not a large carrier power is desired, since either signal can be generated with about the same amount of difficulty. Reference 4 describes a method used to generate all three types of modulations using digital circuitry. The principle behind this method is to generate a square wave version of the modulated signal and then to filter out that portion of the spectrum centered about the harmonics of the carrier frequency. Figure 5 illustrates the output of the modulator, after being filtered by a $Q = 4$ bandpass filter.
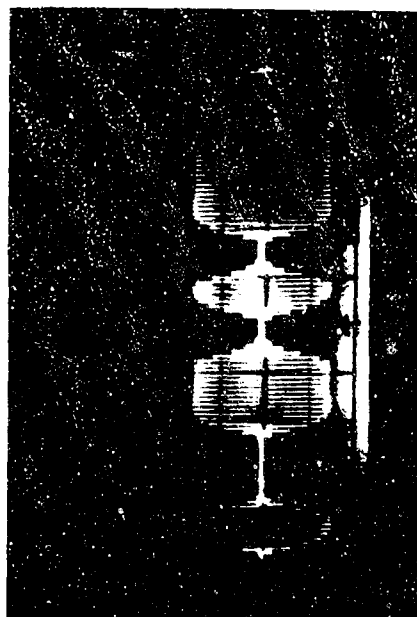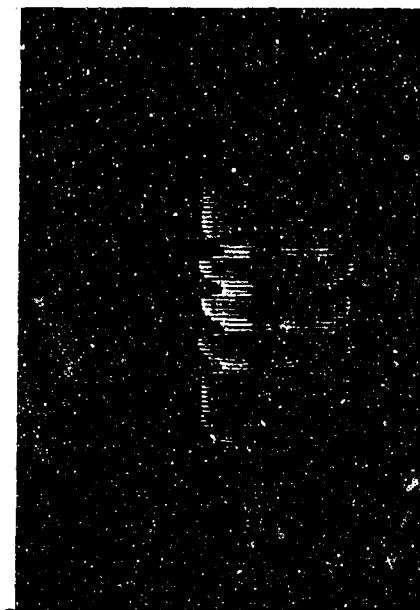
(a)
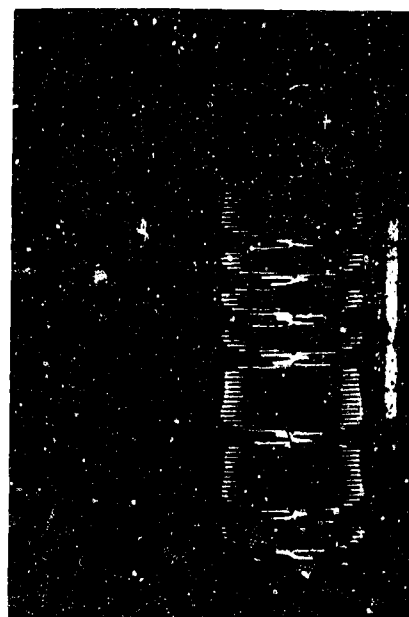
(b)

(c)

Fig. 5. The output of the digital modulator after being filtered by a $Q = 4$ bandpass filter
(a) AM, (b) BM and (c) CM

4. A REPLICA-CORRELATION TECHNIQUE

As mentioned in the introduction, the high resolution property of repeated pulse modulation can be achieved for LMSR sequence modulation by using replica-correlation at the receiver. To be more specific, assume that the transmitted signal can be considered as being propagated over a finite number of distinct transmission paths with transient times $\tau_1, \ldots, \tau_N$ and attenuations $\alpha_1, \ldots, \alpha_N$. We may then write the demodulated reception, $r(t)$, as the sum

$$r(t) = \sum_{i=1}^{N} \alpha_i \; \hat{s}(t - \tau_i) \tag{21}$$

where $\hat{s}(\cdot)$ is the low-pass signal associated with the transmission. Now a standard technique for processing a repeated pulse transmission is to cross-correlate $r(t)$ with a replica of the pulse envelope to produce the function

$$C(\tau) = \sum_{i=1}^{N} \alpha_i \; R(\tau - \tau_i) \tag{22}$$

where $R(\cdot)$ is a triangularly shaped function with base width $2d$. From Eq. 22, it is seen that the ith transmission path appears in $C(\tau)$ as the triangular peak $R(\tau - \tau_i)$ centered at the transient time $\tau_i \pmod{T}$ with height proportional to the attenuation α_i. Thus, $C(\tau)$ may be used as a measure of the multipath structure of the

-17-

channel. In the following paragraphs, a replica-correlation technique is described for obtaining $C(\tau)$ when $\hat{s}(t)$ is the low-pass waveform associated with LMSR sequence modulation.

To begin, assume that a biphase modulated signal is transmitted, $[\hat{s}(t) = \hat{s}_b(t)]$, and consider the cross-correlation of $r(t)$ with the low-pass signal $\hat{s}_a(t)$. From Eq. 21, we have

$$R_{a,r}(\tau) = \frac{1}{T} \int_0^T \hat{s}_a{}^*(t)\, r(t+\tau)\, dt$$

$$= \frac{1}{T} \int_0^T \hat{s}_a{}^*(t) \left[\sum_{i=1}^N \alpha_i\, \hat{s}_b(t-\tau_i) \right] dt$$

$$= \sum_{i=1}^N \alpha_i \left[\frac{1}{T} \int_0^T \hat{s}_a{}^*(t+\tau-\tau_i)\, dt \right]$$

or

$$R_{a,r}(\tau) = \sum_{i=1}^N \alpha_i\, R_{a,b}(\tau-\tau_i) \tag{23}$$

where

$$R_{a,r}(\tau) = \frac{1}{T} \int_0^T \hat{s}_a{}^*(t)\, s_b(t+\tau)\, dt \tag{24}$$

But by substituting for $\hat{s}_a(t)$ and $\hat{s}_b(t)$ from Eq. 8 into Eq. 24 it is easily seen that for $|\tau| \leq T/2$

$$R_{a,b}(\tau) = \begin{cases} (1 + 1/L) \left(1 - \dfrac{|\tau|}{d}\right) & |\tau| \leq d \\ \\ 0 & \text{otherwise} \end{cases} \qquad (25)$$

Thus, $R_{a,b}(\tau)$ is a triangularly shaped function with base width $2d$ and height $1 + 1/L$. [Note that the graph of $R_{a,b}(\tau)$ can be obtained by adding $1/L$ to the graph of $R_b(\tau)$ in Fig. 2(a).] It then follows from Eq. 23 that if a biphase modulated signal is transmitted, we may determine $C(\tau)$ as

$$C(\tau) = R_{a,r}(\tau) \qquad (26)$$

Next, consider the result of transmitting either an amplitude-modulated signal or a complement-phase modulated signal and again cross-correlating the demodulated reception with a replica of $\hat{s}_a(t)$. If the low-pass transmitted signal is denoted by $\hat{s}(t)$, then by proceeding as before, we have,

$$R_{a,r}(\tau) = \sum_{i=1}^{N} \alpha_i R_{a,\hat{s}}(\tau - \tau_i) \qquad (27)$$

where

$$R_{a,\hat{s}}(\tau) = \frac{1}{T} \int_0^T \hat{s}_a^*(t) \, \hat{s}(t + \tau) \, dt \qquad (28)$$

In the Appendix it is shown that $R_{a,\hat{s}}(\tau)$ can be written as

$$R_{a,\hat{s}}(\tau) = C_1 R_{a,b}(\tau) + C_2 \qquad (29)$$

where C_1 and C_2 are constants depending on whether $\hat{s}(t)$ is an AM signal or a CM signal. Thus, we may substitute for $R_{a,\hat{s}}(\tau)$ from Eq. 29 into Eq. 27 to obtain

$$R_{a,r}(\tau) = \sum_{i=1}^{N} \alpha_i [C_1 R_{a,b}(\tau - \tau_i) + C_2]$$

or

$$R_{a,r}(\tau) = C_1 \left[\sum_{i=1}^{N} \alpha_i R_{a,b}(\tau - \tau_i) \right] + C_2 \left[\sum_{i=1}^{N} \alpha_i \right] \qquad (30)$$

Next note that the first bracketed term on the right-hand side of Eq. 30 is equal to $C(\tau)$ so that Eq. 30 may be rearranged to give,

$$C(\tau) = C_1^{-1} \left\{ R_{a,r}(\tau) - C_2 \left[\sum_{i=1}^{N} \alpha_i \right] \right\} \qquad (31)$$

It remains to find an expression for

$$\sum_{i=1}^{N} \alpha_i$$

in terms of a measurable quantity. To this end, consider the mean of the received signal

$$R_{1,r}(\tau) = \frac{1}{T} \int_0^T r(t + \tau)\, dt$$

$$= \frac{1}{T} \int_0^T \left[\sum_{i=1}^N \alpha_i \, \hat{s}(t + \tau - \tau_i) \right] dt$$

$$= \sum_{i=1}^N \alpha_i \frac{1}{T} \int_0^T \hat{s}(t + \tau - \tau_i)\, dt$$

or, since $\hat{s}(t)$ is periodic with period T,

$$R_{1,r}(\tau) = R_{1,r} = \left[\sum_{i=1}^N \alpha_i \right] \left[\frac{1}{T} \int_0^T \hat{s}(t)\, dt \right] \tag{32}$$

Finally, substituting for $\left[\sum_{i=1}^N \alpha_i \right]$ from Eq. 32 into Eq. 31 we obtain

$$C(\tau) = A\, R_{a,r}(\tau) - B\, R_{1,r} \tag{33}$$

where

$$A = C_1^{-1}$$

$$B = [C_2/C_1] \left[\frac{1}{T} \int_0^T \hat{s}(t)\, dt \right]^{-1}$$

From Eqs. 33 and 26 it is seen that $C(\tau)$ can be determined as a linear combination of $R_{a,r}(\tau)$ and $R_{1,r}$ for each of the three modulations considered in this report. The advantage of this technique is that only the constants A and B depend on which modulation is used and thus the basic form of the receiver does not have to

be changed when the modulation is changed.

The constants A and B for the three modulations are derived in the Appendix and summarized in Table III below.

Table III. The coefficients A and B

	A	B
AM	2	-j
BM	1	0
CM	$\sqrt{2}$	$(-j/\sqrt{2}) \dfrac{[1+\frac{1}{L}]}{[1+j/L]} \approx -j/\sqrt{2}$

APPENDIX

In this appendix the coefficients in Table III are derived. To this end it is first necessary to derive the coefficients in Eq. 29. We begin by noting from Eqs. 5 and 7 that

$$\hat{s}_a(t) = [j + \hat{s}_b(t)]/2 \tag{34}$$

and

$$\hat{s}_c(t) = [1 + \hat{s}_b(t)]/\sqrt{2} \tag{35}$$

For the AM case, we set $\hat{s}(t) = \hat{s}_a(t)$ in Eq. 28 and then substitute for $\hat{s}_a(t)$ from Eq. 34 to give

$$R_{a,a}(\tau) = \frac{1}{T} \int_0^T \hat{s}_a{}^*(t)\, \hat{s}_a(t + \tau)\, dt$$

$$= \frac{1}{T} \int_0^T \hat{s}_a{}^*(t) \left[\frac{j + \hat{s}_b(t + \tau)}{2} \right] dt$$

or

$$R_{a,a}(\tau) = \frac{(j)}{2} \left[\frac{1}{T} \int_0^T \hat{s}_a{}^*(t)\, dt \right] + \frac{1}{2}\, R_{a,b}(\tau) \tag{36}$$

Thus, by comparing Eq. 36 with Eq. 29 it is seen that for the AM case

$$C_1 = \frac{1}{2} \tag{37}$$

-23-

and

$$C_2 = (j/2) \left[\frac{1}{T} \int_0^T \hat{s}_a{}^*(t)\, dt \right] \tag{38}$$

For the CM case, we set $\hat{s}(t) = \hat{s}_c(t)$ in Eq. 28 and then substitute for $\hat{s}_c(t)$ from Eq. 35 yielding

$$R_{a,c}(\tau) = \frac{1}{T} \int_0^T \hat{s}_a{}^*(t)\, \hat{s}_c(t + \tau)\, dt$$

$$= \frac{1}{T} \int_0^T \hat{s}_a{}^*(t)\, \frac{[1 + \hat{s}_b(t + \tau)]}{\sqrt{2}}\, dt$$

or

$$R_{a,c}(\tau) = \frac{1}{\sqrt{2}} \left[\frac{1}{T} \int_0^T \hat{s}_a{}^*(t)\, dt \right] + \frac{1}{\sqrt{2}}\, R_{a,b}(\tau)$$

Thus, for the CM case,

$$C_1 = \frac{1}{\sqrt{2}} \tag{39}$$

and

$$C_2 = \frac{1}{\sqrt{2}} \left[\frac{1}{T} \int_0^T \hat{s}_a{}^*(t)\, dt \right] \tag{40}$$

The next step in determining the constants A and B is to compute the means of the low-pass signals $\hat{s}_a(t)$ and $\hat{s}_c(t)$. From Eq. 5,

$$\frac{1}{T} \int_0^T \hat{s}_a(t)\, dt = j \left[\frac{1}{T} \int_0^T \left[\frac{1 + m(t)}{2} \right] dt \right] \tag{41}$$

or

$$\frac{1}{T} \int_0^T \hat{s}_a(t)\, dt = j \left[\frac{L + 1}{2} \right] \frac{1}{L} \tag{42}$$

where we have used Property 2 of Chapter 2 to evaluate the integral on the right-hand side of Eq. 41.

In a similar manner it can be shown that

$$\frac{1}{T} \int_0^T \hat{s}_c(t)\, dt = \frac{1}{T} \int_0^T \left[\frac{1 + j\, m(t)}{\sqrt{2}} \right] dt$$

or

$$\frac{1}{T} \int_0^T \hat{s}_c(t)\, dt = [1 + (j/L)] \frac{1}{\sqrt{2}} \tag{43}$$

Finally, the constant A_1 is determined by substituting for C_1 from Eqs. 37 and 39 into $A = 1/C_1$. The constant B, in the AM case, is given by

$$B = \frac{C_2}{C_1} \left[\frac{1}{T} \int_0^T \hat{s}_a(t)\, dt \right]^{-1}$$

$$= \frac{(j/2)}{(1/2)} \left[\frac{\dfrac{1}{T} \int_0^T \hat{s}_a^*(t)\, dt}{\dfrac{1}{T} \int_0^T \hat{s}_a(t)\, dt} \right] \tag{44}$$

$$= -j$$

and, in the CM case, by

$$B = \frac{C_2}{C_1} \left[\frac{1}{T} \int_0^T \hat{s}_c(t)\, dt \right]^{-1}$$

$$= \frac{(1/\sqrt{2})}{(1/\sqrt{2})} \frac{\left[\frac{1}{T} \int_0^T \hat{s}_a{}^*(t)\, dt \right]}{\left[\frac{1}{T} \int_0^T \hat{s}_c(t)\, dt \right]} \tag{45}$$

$$= \frac{(-j/2)\,(1 + 1/L)}{(1/\sqrt{2})(1 + j/L)}$$

$$= (-j/\sqrt{2}) \left[\frac{1 + 1/L}{1 + j/L} \right]$$

REFERENCES

1. T. G. Birdsall, "MIMI Processing Techniques," (Internal Memorandum), Cooley Electronics Laboratory, The University of Michigan, Ann Arbor, Michigan, December 1968.

2. S. W. Golomb, Shift Register Sequences, Chapters III and IV, Holden-Day, Inc.

3. C. C. Hoopes and R. Randall, Study of Linear Sequence Generators, Cooley Electronics Laboratory Technical Report No. 165 (6576-4-T), The University of Michigan, Ann Arbor, Michigan, June 1966.

4. J. Stewart, "Operating Principles of the ELMSG," (Internal Memorandum), Cooley Electronics Laboratory, The University of Michigan, Ann Arbor, Michigan, July 1971.